Congestion Control in TCP

Antonio Carzaniga

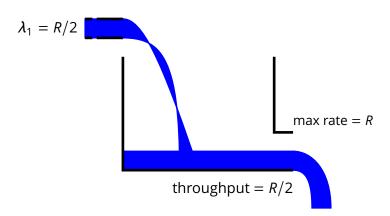
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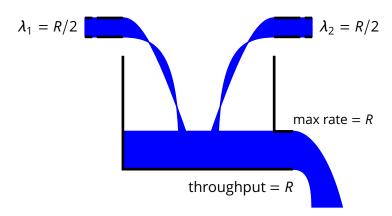
April 20, 2020

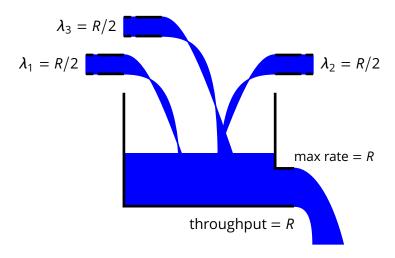
Outline

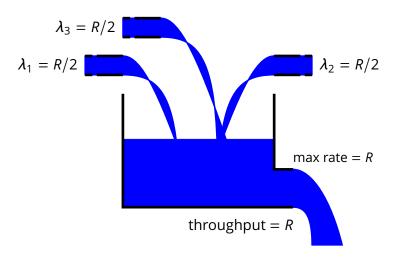
- Intro to congestion control
- Input rate vs. output throughput
- Congestion window
- "Congestion avoidance"
- "Slow start"
- "Fast recovery"

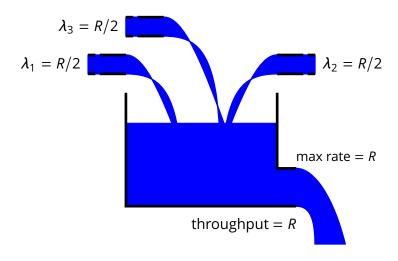


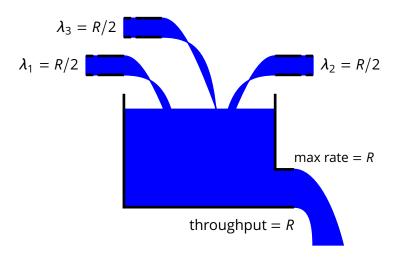


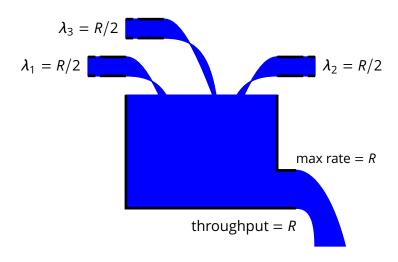














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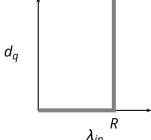


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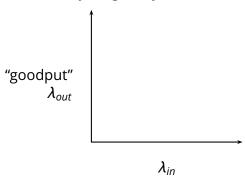
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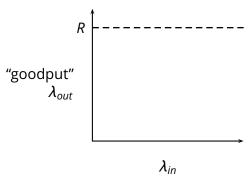


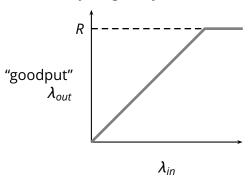
simplified model (D/D/1)

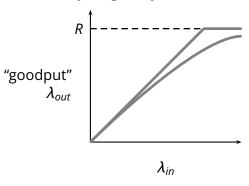
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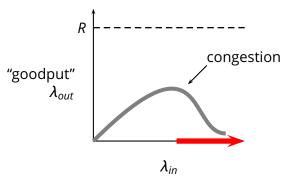






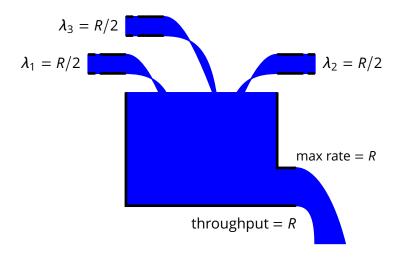


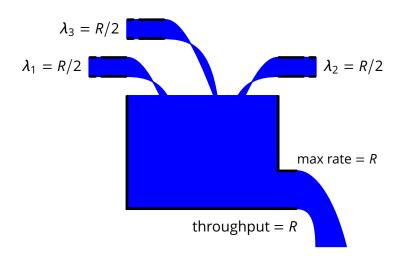


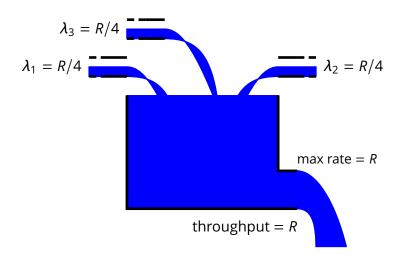


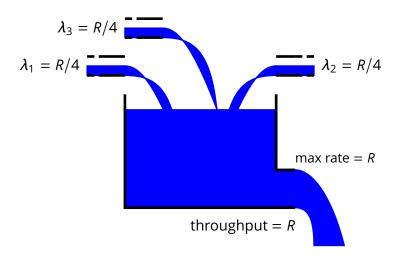
- More realistic assumptions and models
 - finite queue length (buffers) in routers
 - effects of retransmission overhead
 - full queues along multi-hops paths

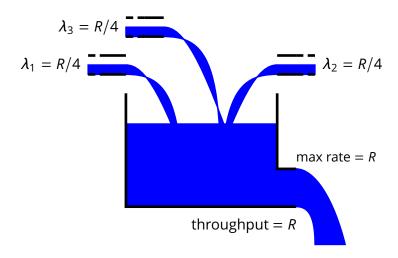
■ What to do when the network is congested?



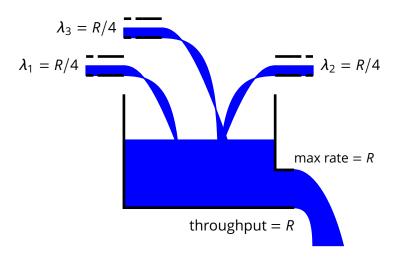








■ What to do when the network is congested? **BACK OFF!**



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- 3. how should the sender *control its output rate*?
 - we need a brain and we need to know how to drive!



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- Congestion means that some queues overflow in one or more routers between the sender and the receiver
 - the visible effect is that some segments are dropped
- Therefore the sender assumes that the network is congested when it (the sender) detects a segment loss
 - duplicate acknowledgements (i.e., NACK)
 - time out (i.e., no ACKs at all)

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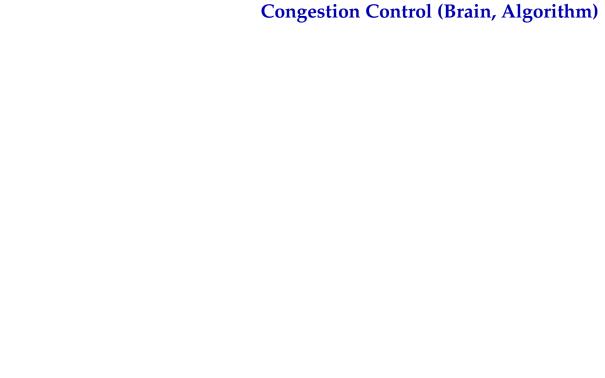
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■ The resulting maximum output rate is roughly

$$\lambda = \frac{W}{2L}$$



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- Additive-increase and multiplicative-decrease
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- Reaction to timeout events



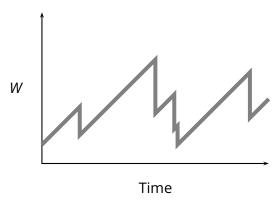
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 - e.g., suppose W = 14600 and MSS = 1460, then the sender increases W to 16060 after 10 acknowledgments

■ Window size W over time



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- This process is called "slow start" because of the small initial value of W

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- So, TCP reacts differently to a timeout and to a triple duplicate ACKs

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 - ▶ then proceed with congestion avoidance
- *NACK* (i.e., triple duplicate-ack)
 - set ssthresh = $\overline{W}/2$
 - cut W in half: $W = \overline{W}/2$
 - ▶ run *congestion avoidance*, ramping up *W* linearly
 - ► This is called *fast recovery*

